AMENDMENTS TO THE CLAIMS

This listing of claims will replace all prior versions, and listings, of claims in the application:

Listing of Claims:

1. (Currently amended) An apparatus that provides a unified telephony solution, comprising:

an application server configured to provide telephony services;

a voice extensible markup language (VXML) browser configured to access telephony services through the application server;

a telephony controller configured to access telephony services through the VXML browser, wherein the telephony controller includes a SIP framework with a SIP servlet container, wherein the SIP servlet container includes a plurality of SIP servlets for interfacing with a SIP network; and

a telephony gateway that provides an interface to a public switched telephone network (PSTN).

2. (Canceled).

3. (Currently amended) The apparatus of ~~claim 2~~ claim 1, wherein a new telephony service can be added by including a new SIP servlet in the SIP servlet container.

4. (Currently amended) The apparatus of ~~claim 2~~ claim 1, wherein the plurality of SIP servlets are registered with a remote method invocation (RMI) registry.

5. (Currently amended) The apparatus of ~~claim 2~~ claim 1, wherein the telephony services provide at least two of:

a telephone system;

a call center;

an interactive voice response (IVR) system; and

a voicemail system.

6. (Currently amended) The apparatus of ~~claim 2~~ claim 1, wherein the apparatus operates using a Voice Over Internet Protocol (VOIP).

7. (Currently amended) The apparatus of ~~claim 2~~ claim 1, wherein the application server is coupled to a database that provides access to the plurality of SIP servlets.

8. (Currently amended) A method that provides a unified telephony solution, comprising:

receiving a request for a telephony service at a telephony controller, wherein the telephony controller includes a SIP framework with a SIP servlet container, wherein the SIP servlet container includes a plurality of SIP servlets for interfacing with a SIP network; and

in response to the request, accessing a telephony service provided by an application server;

wherein the application server is accessed through a voice extensible markup language (VXML) browser;

wherein performing the telephony service involves interfacing to a public switched telephone network (PSTN) through a telephony gateway.

9. (Canceled)

10. (Currently amended) The method of ~~claim 9~~ claim 8, wherein a new telephony service can be added by including a new SIP servlet in the SIP servlet container.

11. (Currently amended) The method of ~~claim 9~~ claim 8, wherein the plurality of SIP servlets are registered with a remote method invocation (RMI) registry.

12. (Currently amended) The method of ~~claim 9~~ claim 8, wherein the telephony services provide at least two of:

a telephone system;

a call center;

an interactive voice response (IVR) system; and

a voicemail system.

13. (Currently amended) The method of ~~claim 9~~ claim 8, wherein the telephony services operate using the Voice Over Internet Protocol (VOIP).

14. (Currently amended) The method of ~~claim 9~~ claim 8, wherein the application server is coupled to a database that provides access to the plurality of SIP servlets.

15. (Currently amended) A computer-readable storage ~~medium~~device storing instructions that when executed by a computer cause the computer to perform a method that provides a unified telephony solution, the method comprising:

receiving a request for a telephony service at a telephony controller, wherein the telephony controller includes a SIP framework with a SIP servlet container, wherein the SIP servlet container includes a plurality of SIP servlets for interfacing with a SIP network; and

in response to the request, accessing a telephony service provided by an application server through a voice extensible markup language (VXML) browser;

wherein the telephony service involves interfacing to a public switched telephone network (PSTN) through a telephony gateway.

16. (Canceled)

17. (Currently amended) The computer-readable storage ~~medium~~device of ~~claim 16~~ claim 15, wherein a new telephony service can be added by including a new SIP servlet in the SIP servlet container..

18. (Currently amended) The computer-readable storage ~~medium~~device of ~~claim 16~~ claim 15, wherein the plurality of SIP servlets are registered with a remote method invocation (RMI) registry.

19. (Currently amended) The computer-readable storage ~~medium~~device of ~~claim 16~~ claim 15, wherein the telephony services provide at least two of:

a telephone system;

a call center;

an interactive voice response (IVR) system; and

a voicemail system.

20. (Currently amended) The computer-readable storage ~~medium~~device of ~~claim 16~~ claim 15, wherein the telephony services operate using the Voice Over Internet Protocol (VOIP).

21. (Currently amended) The computer-readable storage ~~medium~~device of ~~claim 16~~claim 15, wherein the application server is coupled to a database that provides access to the plurality of SIP servlets.